

IN THE CLAIMS:

Please amend the claims as follows:

1. (previously presented) A method, comprising:
segmenting an audio signal into a plurality of segments based on audio characteristics of the audio signal, the audio characteristics indicative of parameters in a parametric representation of the audio signal; and
encoding the segments with different encoding settings.
2. (canceled)
3. (original) The method of claim 1, wherein the characteristics include voicing characteristics in said segments of the audio signal.
4. (previously presented) The method of claim 1, wherein the characteristics include energy characteristics in said segments of the audio signal.
5. (previously presented) The method of claim 1, wherein the characteristics include pitch characteristics in said segments of the audio signal.
6. (previously presented) The method of claim 1, wherein said segmenting is carried out concurrent to said encoding.
7. (previously presented) The method of claim 1, wherein said segmenting is carried out before said encoding.
8. (previously presented) The method of claim 1, wherein a plurality of voicing values are assigned to the audio characteristics of the audio signal in said segments, and wherein said segmenting is carried out based on the assigned voicing values.

9. (original) The method of claim 8, wherein the plurality of values includes a value designated to a voiced speech signal and another value designated to an unvoiced signal.

10. (original) The method of claim 8, wherein the plurality of values further includes a value designated to a transitional stage between the voice and unvoiced signal.

11. (previously presented) The method of claim 8, wherein the plurality of values further includes a value designated to an inactive period in the audio signal.

12. (previously presented) The method of claim 1, further comprising selecting a quantization mode for said encoding in order to improve the bit allocation and to reduce the parameter update rate, wherein the segmenting is carried out based on the selected quantization mode.

13. (previously presented) The method of claim 1, wherein said segmenting is carried out based on a selected target accuracy in reconstructing of the audio signal, wherein the target accuracy is selected based on a distortion criteria comparing upsampled quantized values and modified parameter signal.

14. (previously presented) The method of claim 5, wherein said segmenting is carried out for providing a linear pitch representation in at least some of said segments.

15. (previously presented) The method of claim 1, wherein the audio signal is encoded into audio signal data, said method further comprising:

forming a parameter signal based on the audio signal data having a first number of signal data;

downsampling the parameter signal to a second number of signal data for providing a further parameter signal, wherein the second number is smaller than the first number; and

upsampling the further parameter signal to a third number of signal data in decoding, wherein the third number is greater than the second number.

16. (original) The method of claim 15, wherein the third number is equal to the first number.

17. (original) The method of claim 15, wherein the signal data comprise quantized parameters.

18. (original) The method of claim 15, wherein the signal data comprises unquantized parameters.

19. (previously presented) A decoder, comprising:

an input for receiving audio data indicative of a plurality of parameters in an adjusted representation, wherein the audio data comprises a plurality of segments indicative of an input audio signal having audio characteristics and wherein the segments are obtained based on the audio characteristics and encoded with a plurality of encoding settings based on the audio characteristics; and

a module, responsive to the audio data, for generating a further audio signal based on the adjusted representation and the encoding settings.

20. (original) The decoder of claim 19, wherein the audio data is recorded on an electronic medium, and wherein input of the decoder is operatively connected to the electronic medium for receiving the audio data.

21. (original) The decoder of claim 19, wherein the audio data is transmitted through a communication channel, and wherein the input of the decoder is operatively connected to the communication channel for receiving the audio data.

22. (previously presented) An encoding device comprising:

an input for receiving audio data indicative of audio characteristics; and

an adjustment module for adjusting one or more parameters based on the audio characteristics for providing an adjusted representation of the parameters, wherein said adjusting comprises segmenting the audio signal into a plurality of segments based on the characteristics of the audio signals and encoding the segments based on one or more of a plurality of encoding settings.

23. (previously presented) The encoding device of claim 22, further comprising a quantization module, responsive to the adjusted representation, for coding the parameters in the adjusted representation.

24. (previously presented) The encoding device of claim 22, further comprising an output end, operatively connected to a storage medium, for providing data indicative of the coded parameters in the adjusted representation to the storage medium for storage.

25. (previously presented) The encoding device of claim 22, further comprising an output end, operatively connected to a communication channel, for providing signals indicative of the coded parameters in the adjusted representation to the communication channel for transmission.

26. (currently amended) A computer readable storage medium ~~having an embedded with a computer program software product, said computer software product~~ comprising programming codes for carrying out the method of claim 1.

27. (previously presented) An electronic device comprising:

an input module for receiving audio data indicative of a plurality of parameters in an adjusted representation, wherein the audio data comprises a plurality of segments indicative of an input audio signal having audio characteristics and wherein the segments are obtained based on the audio characteristics and encoded with a plurality of encoding settings based on the audio characteristics; and

a decoder, responsive to the audio data, for generating a synthesized audio signal based on the adjusted representation.

28. (original) The electronic device of claim 27, wherein the audio data is recorded in an electronic medium, and wherein the input is operatively connected to the electronic medium for receiving the audio data.

29. (original) The electronic device of claim 27, wherein the audio data is conveyed through a communication channel, and wherein the input is operatively connected to the communication channel for receiving the audio data.

30. (original) The electronic device of claim 27, comprises a mobile terminal.

31. (previously presented) A communication network, comprising:

- a plurality of base stations; and

- a plurality of mobile stations adapted for communicating with the base stations, wherein at least one of the mobile stations comprises:

- an input module for receiving audio data indicative of a plurality of parameters in an adjusted representation from at least one of the base stations, wherein the audio data comprises a plurality of segments indicative of an input audio signal having audio characteristics and wherein the segments are obtained based on the audio characteristics and encoded with a plurality of encoding settings based on the audio characteristics; and

- a decoder, responsive to the audio data, for generating a synthesized audio signal based on the adjusted representation.

32. (previously presented) A decoder, comprising:

- an input for receiving audio data indicative of end points defining a plurality of sub-segments, wherein the audio signal is indicative of a plurality of parameters, the parameters including pitch contour data containing a plurality of pitch values representative of an audio segment in time, and wherein the pitch contour data in the audio segment in time is approximated by a plurality of consecutive sub-segments in the audio segment, and wherein the end points include a first end point and a second end point for defining each of said sub-segments; and

- a reconstruction module for reconstructing the audio segment based on the received audio data.

33. (previously presented) The method of claim 1, wherein the encoding settings include bit allocation, quantization accuracy, quantization method and parameter update rate.

34. (previously presented) The method of claim 1, wherein the audio signal contains sinusoidal components and said parameters include frequency values, amplitude values and phase values indicative of the sinusoidal components.
35. (previously presented) The method of claim 1, wherein the parameters include pitch, voicing, amplitude and energy of the audio signal.
36. (previously presented) The method of claim 1, wherein the parameters include pitch contour data containing a plurality of pitch values representative of an audio segment in time.
37. (previously presented) The decoder of claim 19, wherein the encoding settings include bit allocation, quantization accuracy, quantization method and parameter update rate.
38. (previously presented) The encoding device of claim 22, wherein the encoding settings include bit allocation, quantization accuracy, quantization method and parameter update rate.
39. (currently amended) The computer readable storage medium embedded with a computer program of claim 26, wherein the encoding settings include bit allocation, quantization accuracy, quantization method and parameter update rate.
40. (previously presented) The communication network of claim 31, wherein the encoding settings include bit allocation, quantization accuracy, quantization method and parameter update rate.
41. (previously presented) The method of claim 1, wherein the audio signal comprises a plurality of frames and the audio signal in each frame has a waveform and wherein a further audio signal is produced in the decoding stage independently of the waveform.

42. (previously presented) The method of claim 1, wherein each segment has a segment length and wherein the segment length of at least one segment is different from the segment length of at least one other segment.

43. (previously presented) The decoder of claim 19, wherein the audio signal comprises a plurality of frames and the audio signal in each frame has a waveform and wherein the module generates the further audio signal independently of the waveform.

44. (previously presented) The decoder of claim 19, wherein the segments comprise segments of different segment lengths.

45. (previously presented) The encoding device of claim 22, wherein the segments comprise segments of different segment lengths.

46. (previously presented) The electronic device of claim 26, wherein the segments comprise segments of different segment lengths.

47. (previously presented) The communication network of claim 31, wherein the segmented and encoded segments comprise segments of different segment lengths.

48. (previously presented) The decoder of claim 32, wherein the sub-segments comprise sub-segments of different segment lengths.